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Blind Audio Source Separation on Tensor Representation



ie University of Tokyo

Hiroshi Sawada (NTT Corporation) Nobutaka Ono (Tokyo Metropolitan University) Hirokazu Kameoka (NTT Corporation) Daichi Kitamura (The University of Tokyo)

Tutorial structure

1. Introduction

- 1. Separation of audio/speech signals
- 2. Live demonstration

2. ICA and IVA

- 1. ICA: Independent Component Analysis
- 2. IVA: Independent Vector Analysis

3. NMF

- 1. NMF: Nonnegative Matrix Factorization
- 2. MNMF: Multichannel NMF

4. ILRMA

1. ILRMA: Independent Low-Rank Matrix Analysis

Separation of audio/speech signal

cocktail party effect



speech recognition in noisy environment





Live demonstration

- Separate 2 speeches with 2 microphones
- iPhone app



• Script

The ICASSP meeting is the world's largest and most comprehensive technical conference focused on signal processing and its applications. We are demonstrating the Blind Source Separation for convolutive mixtures of speech in a real room with real talkers.

BSS: Blind Source Separation

- Separate the mixtures at microphones x_1, x_2
 - into the original sources s_1, s_2
 - assuming M=N=2 for simplicity
 - source activity and mixing system H is unknown (blind)



Instantaneous BSS

- Mixing system **H** is described by scalars
- Sources are multiplied by scalars and then mixed



Convolutive BSS

Delay and reverberations in a real room situation
 mixing system H is described by impulse responses

Sources are convolutively mixed



 $x_m(t) = \sum_{n=1}^{N} \sum_{\tau=0}^{L-1} h_{mn}(\tau) s_n(t-\tau)$

Convolutive BSS is a much harder problem than instantaneous BSS

$$x_m(t) = \sum_{n=1}^N h_{mn} s_n(t)$$

The whole system



STFT: short-time Fourier transform

- From a time-domain real-valued signal
- To a time-frequency-domain complex-valued signal



Three key axes

Frequency

- Frequency-domain processing is effective
 - source characteristic
 - convolution \rightarrow multiplication

Time

Source activity, Onset and offset

Channel

• Source, Mixture, Separation

Tensor representation



Tensor and sliced matrices



Notations

- s: sources
- **x**: mixtures/observations
- y: separations
- W: separation system H: mixing system
 - \mathbf{H}_i mixing matrix
 - \mathbf{h}_{in} mixing vector
 - H_{in} spatial covariance

- *i*: frequency bin index*j*: time frame index*m*: microphone index*n*: source/separation index
- *I*: number of frequency bins*J*: number of time frames*M*: number of microphones*N*: number of sources

- **T**: basis spectrum
- V: time-varying magnitude (NMF), whitening matrix (FastICA), weighted covariance matrix (AuxIVA)

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Tensor and sliced matrices



ICA: Independent Component Analysis

- Extract the original sources from the mixtures x_1, x_2
- Mixing matrix **H** cannot be obtained



Sources are assumed to be independent

Makes y_1, y_2 independent to each other

ICA: Independent Component Analysis

In addition to independence, need to assume
 source distributions are different from Gaussian



Distributions are different from Gaussian



Makes the distribution far from Gaussian

Speech mixtures





Statistics of the mixtures



Statistics of the mixtures



A complex-valued source model

Super-Gaussian distribution

$$p(y_n) \propto \exp\left(-\frac{\sqrt{|y_n|^2 + \alpha}}{b}\right) \qquad \qquad y_n \in \mathbb{C}$$

• Sharper peak at the origin than Gaussian



Independent component analysis



- $\mathbf{y}(j) = \mathbf{W}\mathbf{x}(j)$ Linear operation
- Output independence $p(\mathbf{y}) = \prod_{n=1}^{N} p(y_n)$
- Non-gaussianity

$$p(y_n) \neq \frac{1}{\pi \sigma^2} \exp\left(-\frac{|y_n|^2}{\sigma^2}\right)$$

$$\mathbf{x} = egin{bmatrix} x_1 \ dots \ x_M \end{bmatrix} \ \mathbf{y} = egin{bmatrix} y_1 \ dots \ y_N \end{bmatrix}$$

 $\mathbf{x} =$

Likelihood of separation matrix W

- Likelihood of W for the whole observations $p(\mathcal{X}|\mathbf{W}) = \prod_{j=1}^{J} p(\mathbf{x}(j)|\mathbf{W})$ $\mathcal{X} = \{\mathbf{x}(1), \dots, \mathbf{x}(J)\}$
- Probability density function, linear transformation $p(\mathbf{x}|\mathbf{W}) = |\det \mathbf{W}|^2 p(\mathbf{y})$ $\longleftarrow \mathbf{y}(j) = \mathbf{W}\mathbf{x}(j)$
- Output independence $p(\mathbf{y}) = \prod_{n=1}^{N} p(y_n)$

Log-likelihood function

 $\log p(\mathcal{X}|\mathbf{W}) = 2J \cdot \log |\det \mathbf{W}| + \sum_{j=1}^{J} \sum_{n=1}^{N} \log p(y_n(j))$

Objective function to be minimized

Maximum likelihood estimation = Minimize the negative log-likelihood $\mathcal{J}(\mathbf{W}) = J\left[\sum_{n=1}^{N} E\left\{G(y_n)\right\} - 2\log|\det \mathbf{W}|\right]$

Contrast function

$$G(y_n) = -\log p(y_n)$$

1st order derivative

$$g(y_n) = \frac{\partial G(y_n)}{\partial y_n^*} = -\frac{\partial \log p(y_n)}{\partial y_n^*}$$

2nd order derivative

$$g'(y_n) = \frac{\partial^2 G}{\partial y_n^* \partial y_n} = \frac{\partial g}{\partial y_n}$$

 $G(y_n) = \sqrt{|y_n|^2 + \alpha}$

$$g(y_n) = \frac{y_n}{2\sqrt{|y_n|^2 + \alpha}}$$

$$p(y_n) \propto \exp\left(-\frac{\sqrt{|y_n|^2 + \alpha}}{b}\right)$$
(b) Assumed, $\alpha = 0.1$
(c) As

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$$g'(y_n) = \frac{1}{2\sqrt{|y_n|^2 + \alpha}} \left[1 - \frac{1}{2} \frac{|y_n|^2}{|y_n|^2 + \alpha} \right]$$

Optimization methods

- Gradient descent $\begin{aligned}
 \frac{\partial 2 \log |\det \mathbf{W}|}{\partial \mathbf{W}^*} &= (\mathbf{W}^{\mathsf{H}})^{-1} \\
 \mathbf{W} \leftarrow \mathbf{W} - \eta \cdot \frac{1}{J} \frac{\partial \mathcal{J}}{\partial \mathbf{W}^*} & \eta \text{ :step size} \\
 \frac{1}{J} \frac{\partial \mathcal{J}}{\partial \mathbf{W}^*} &= \mathrm{E}\{\mathbf{g}(\mathbf{y})\mathbf{x}^{\mathsf{H}}\} - (\mathbf{W}^{\mathsf{H}})^{-1} & \mathbf{g}(\mathbf{y}) = \begin{bmatrix} g(y_1) \\ \vdots \\ g(y_N) \end{bmatrix}
 \end{aligned}$
 - Expensive matrix inversion
 - Slow convergence
- Three practical ways
 - 1. Natural gradient
 - 2. Pre-whitening + FastICA
 - 3. Auxiliary function-based optimization

 $g(y_n) = \frac{y_n}{2\sqrt{|y_n|^2 + \alpha}}$

1. Natural gradient

[Amari et al., 1996]

 $g(y_n) = \frac{y_n}{2\sqrt{|y_n|^2 + \alpha}}$

$$\frac{1}{J} \frac{\partial \mathcal{J}}{\partial \mathbf{W}^*} \mathbf{W}^{\mathsf{H}} \mathbf{W} = \left[\mathrm{E} \{ \mathbf{g}(\mathbf{y}) \mathbf{y}^{\mathsf{H}} \} - \mathbf{I} \right] \cdot \mathbf{W}$$

- No matrix inversion
 - Efficient computation

 $\mathbf{W} \leftarrow \mathbf{W} - \eta \cdot \frac{1}{\overline{\overline{\mathbf{U}}}} \cdot \frac{\partial \mathcal{J}}{\partial \overline{\mathbf{U}}} \mathbf{W}^{\mathsf{H}} \mathbf{W}$

- Equivariance property [Cardoso and Souloumiac, 1996]
 - Free from the characteristics of mixing matrix (e.g. close to singular) $E\{g(y)x^H\}$

Convergence example



2. Pre-whitening + FastICA

Separation matrix of the form: $\mathbf{W} = \mathbf{U}\mathbf{V}^{[Hyvarinen et al., 2001]}$



FastICA

• Objective function w.r.t a unitary matrix **U**

$$\mathcal{J}(\mathbf{U}) = J \left[\sum_{n=1}^{N} \mathrm{E}\{G(y_n)\} - 2\log|\det \mathbf{U}| \right]$$
$$= J \sum_{n=1}^{N} \mathrm{E}\{G(y_n)\} \qquad \longleftarrow \quad \det \mathbf{U} = 1$$

- Minimize $E\{G(y_n)\}$ for y_1, y_2, \dots, y_N $G(y_n) = \sqrt{|y_n|^2 + \alpha}$
 - with unitary constraint

$$\mathbf{u}_{n}^{\mathsf{H}}\mathbf{u}_{k} = \begin{cases} 1 & \text{if } n = k \\ 0 & \text{if } n \neq k \end{cases} \qquad \mathbf{U} = [\mathbf{u}_{1}\cdots\mathbf{u}_{N}]^{\mathsf{H}}$$

FastICA algorithm

[Hyvarinen et al., 2001]

- For $n = 1, \ldots, N$ (sequentially)
 - Iterate the followings until convergence

Separated signal calculation

$$y_n(j) = \mathbf{u}_n^{\mathsf{H}} \mathbf{z}(j)$$

Optimization of G by Newton's method

$$\mathbf{u}_{n} \leftarrow \mathrm{E}\{g(y_{n})\mathbf{z}\} - \mathrm{E}\{g'(y_{n})\}\mathbf{u}_{n}$$
$$g(y_{n}) = \frac{y_{n}}{2\sqrt{|y_{n}|^{2} + \alpha}} \quad g'(y_{n}) = \frac{1}{2\sqrt{|y_{n}|^{2} + \alpha}} \left[1 - \frac{1}{2}\frac{|y_{n}|^{2}}{|y_{n}|^{2} + \alpha}\right]$$

Gram-schmidt orthogonalization

$$\mathbf{u}_n \leftarrow \mathbf{u}_n - \sum_{k=1}^{n-1} (\mathbf{u}_k^\mathsf{H} \mathbf{u}_n) \mathbf{u}_k$$

Unit-norm normalization

$$\mathbf{u}_n \leftarrow rac{\mathbf{u}_n}{||\mathbf{u}_n||}$$

FastICA convergence example

- •Red (\Box)
 - Starting from $\mathbf{u}_1 = \begin{bmatrix} 1 & 0 \end{bmatrix}^\mathsf{T}$
 - Likelihood maximization: points close to the origin
 - Unit-norm normalization: points on the unit sphere
 - Good solution only by 5 iterations
- Green (\triangle)
 - Starting from $\mathbf{u}_2 = \begin{bmatrix} 0 & 1 \end{bmatrix}^\mathsf{T}$
 - One-step solution by orthogonalization



3. Auxiliary function

[Ono and Miyabe, 2010] Objective function to be minimized [Ono 2011] $\mathcal{J}(\mathbf{W}) = J \left| \sum_{n=1}^{N} \mathbb{E} \left\{ G(|y_n|) \right\} - 2 \log |\det \mathbf{W}| \right|$ $G(y_n) = \sqrt{|y_n|^2 + \alpha}$ • If $G(y_n) = G_R(|y_n|)$ and $G_R(|y_n|) = \sqrt{|y_n|^2 + \alpha}$ $\frac{G'_R(|y_n|)}{|y_n|}$ is monotonically decreasing $G'_R(|y_n|) = \frac{\partial G_R}{\partial |y_n|} = \frac{|y_n|}{\sqrt{|y_n|^2 + \alpha}}$ $\frac{G_R'(|y_n|)}{|y_n|} = \frac{1}{\sqrt{|y_n|^2 + \alpha}}$ Auxiliary function $\frac{G_R'(r_n)}{2r_n} |y_n|^2 + F(r_n)$ $G_R(|y_n|)$ $G(y_n) \le \frac{G'_R(r_n)}{2r} |y_n|^2 + F(r_n)$ 3 with auxiliary variable r_n 2 Equal when $r_n = |y_n|$ 1 $r_n = 1$ y_n

0

Auxiliary function-based optimization

[Ono and Miyabe, 2010] Iterate the followings until convergence [Ono 2011] Separated signal calculation $\mathbf{y} = \mathbf{W}\mathbf{x}$ $\mathbf{y} = \begin{bmatrix} y_1 \\ \vdots \\ y_M \end{bmatrix}$ $\mathbf{W} = \begin{bmatrix} \mathbf{w}_1^H \\ \vdots \\ \mathbf{w}_M^H \end{bmatrix}$ Weighted covariance matrices for each separation $\mathbf{V}_n = \mathbf{E} \left\{ \frac{G_R'(|y_n|)}{2|y_1|} \mathbf{x} \mathbf{x}^{\mathsf{H}} \right\}$ $\frac{G_R(|y_n|)}{|y_n|} = \frac{1}{\sqrt{|y_n|^2 + \alpha}}$ Solve the HEAD problem for $\mathbf{V}_1, \ldots, \mathbf{V}_N$ $\mathbf{w}_{m}^{\mathsf{H}} \mathbf{V}_{n} \mathbf{w}_{n} = \begin{cases} 1 & \text{if } m = n \\ 0 & \text{if } m \neq n \end{cases}$

Auxiliary ICA convergence example

Estimated mixing matrix $\mathbf{A} = \mathbf{W}^{-1}$ Starting from $\mathbf{W} = \begin{bmatrix} 1 & 0 \\ 0 & 1 \end{bmatrix}$





at an early stage

Comparisons of three methods

- measured by signal-to-interference ratio (SIR)
- at frequency 3586 Hz, 201 samples, pre-whitening applied



- Natural gradient: sensitive to step-size η
- ✓ FastICA: fast convergence, slightly limited SIR (unitary constraint)
- ✓ Auxiliary ICA: fast convergence

Permutation and Scaling problem

Ambiguities of ICA solutions

If $\mathbf{y}(j) = \mathbf{W} \mathbf{x}(j)$ is a solution, then

 $\mathbf{y}(j) \leftarrow \mathbf{\Lambda} \, \mathbf{P} \, \mathbf{y}(j)$ is also a solution

for any diagonal Λ and permutation \mathbf{P} matrix

$$\begin{bmatrix} y_1 \\ y_2 \\ y_3 \end{bmatrix} \leftarrow \begin{bmatrix} 3 & 0 & 0 \\ 0 & 0.5 & 0 \\ 0 & 0 & 4 \end{bmatrix} \begin{bmatrix} 0 & 0 & 1 \\ 1 & 0 & 0 \\ 0 & 1 & 0 \end{bmatrix} \begin{bmatrix} y_1 \\ y_2 \\ y_3 \end{bmatrix}$$

Independence of y_1, y_2, y_3 does not change
Permutation and Scaling problem



Solving Permutation and Scaling Problems

Permutation

- Post-processing
 - see e.g. [Sawada et al., 2004], [Sawada et al., 2011]
- Tensor methods (IVA, ILRMA)
 - will be explained in later sections

• Scaling

- Refer to a microphone observation
 - So-called "projection back"
 - via mixing system estimation

[Cardoso 1998] [Murata et al., 2001] [Matsuoka and Nakashima 2001] [Takatani et al., 2004]

Scaling alignment example



reference microphone m = 1

 a_{1n} estimated mixing scalar

Mixing system estimation

Estimated mixing situation ICA result

$$\mathbf{x} = \sum_{n=1}^{N} \mathbf{a}_n y_n = \mathbf{A} \mathbf{y} \quad \mathbf{y} = \mathbf{W} \mathbf{x}$$

$$\mathbf{a}_n = \begin{bmatrix} a_{1n} \\ \vdots \\ a_{Mn} \end{bmatrix} \quad \mathbf{A} = [\mathbf{a}_1, \cdots, \mathbf{a}_N] \quad \mathbf{x} = \begin{bmatrix} x_1 \\ \vdots \\ x_M \end{bmatrix}$$

How to calculate matrix A

 ${\scriptstyle \bullet}$ If ${\rm \, W}\,$ has an inverse

$$\mathbf{A} = \mathbf{W}^{-1}$$

$$ullet$$
 Otherwise ($N < M$)

$$\mathbf{A} = \mathbf{E}\{\mathbf{x}\mathbf{y}^H\}(\mathbf{E}\{\mathbf{y}\mathbf{y}^H\})^{-1}$$

 $\mathbf{A} = \mathbf{W}^+$

- Least-mean-square estimator that minimizes $E\{||\mathbf{x} - \mathbf{A}\mathbf{y}||^2\}$
- Moore-Penrose pseudo inverse

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Two papers at ICA 2006

Independent Vector Analysis: An Extension of ICA to Multivariate Components

Taesu Kim^{1,2}, Torbjørn Eltoft³, and Te-Won Lee¹

¹ Institute for Neural Computation, UCSD, USA {taesu, tewon}@ucsd.edu ² Department of BioSystems, KAIST, Korea ³ Department of Physics, University of Tromsø, Norway torbjorn.eltoft@phys.uit.no

Abstract. In this paper, we solve an ICA problem where both source and observation signals are multivariate, thus, vectorized signals. To derive the algorithm, we define dependence between vectors as Kullback-Leibler divergence between joint probability and the product of marginal probabilities, and propose a vector density model that has a variance dependency within a source vector. The example shows that the algorithm successfully recovers the sources and it does not cause any permutation ambiguities within the sources. Finally, we propose the frequency domain blind source separation (BSS) for convolutive mixtures as an application of IVA, which separates 6 speeches with 6 microphones in a reverberant room environment. Solution of Permutation Problem in Frequency Domain ICA, Using <u>Multivariate</u> Probability Density Functions

Atsuo Hiroe

Intelligent Systems Research Laboratory, Information Technologies Laboratories, Sony Corporation, 6-7-35 Kitashinagawa, Shinagawa-ku, Tokyo 141-0001, Japan Atsuo.Hiroe@jp.sony.com

Abstract. Conventional Independent Component Analysis (ICA) in frequency domain inherently causes the permutation problem. To solve the problem fundamentally, we propose a new framework for separation of the whole spectrograms instead of the conventional binwise separation. Under our framework, a measure of independence is calculated from the whole spectrograms, not individual frequency bins. For the calculation, we introduce some multivariate probability density functions (PDFs) which take a spectrum as arguments. To seek the unmixing matrix that makes spectrograms independent, we demonstrate a gradient-based algorithm using multivariate activation functions derived from the PDFs. Through experiments using real sound data, we have confirmed that our framework is effective to generate permutation-free unmixed results.

Concept of "multivariate source model" was presented in two papers at the same conference independently

Independent vector analysis

• ICA: Sources generate stochastic scalar variables



• IVA: Sources generate stochastic vector variables



Same mixing / separation process as ICA, but multivariate source model

Likelihood of separation matrices W

- Likelihood of all Ws for the whole observations $p(\mathcal{X}|\mathcal{W}) = \prod_{j=1}^{J} p(\mathbf{x}_{j,m}|\mathcal{W}) \begin{array}{l} \mathcal{X} = \{\mathbf{x}_{j,m} | j = 1, \cdots, J, m = 1, \cdots, N\} \\ \mathcal{W} = \{\mathbf{W}_1, \mathbf{W}_2, \cdots, \mathbf{W}_I\} \end{array}$ Probability density function, linear transformation
- $p(\mathbf{x}_{j,1},\cdots,\mathbf{x}_{j,N}|\mathcal{W}) = \prod_{i=1}^{N} |\det \mathbf{W}_i|^2 p(\mathbf{y}_{j,1},\cdots,\mathbf{y}_{j,N})$ Output independence

$$\mathbf{p}(\mathbf{y}_{j,1},\cdots,\mathbf{y}_{j,N}) = \prod_{n=1}^{N} p(\mathbf{y}_{j,n})$$

Log-likelihood function

$$\mathcal{L} = \log p(\mathcal{X}|\mathbf{W})$$

= $2J \sum_{i=1}^{I} \log |\det \mathbf{W}_i| + \sum_{j=1}^{J} \sum_{n=1}^{N} \log p(\mathbf{y}_{j, n})$



 $\mathbf{w}_{ij} = \mathbf{W}_i \mathbf{x}_{ij}$

• A set of demixing matrices to be estimated

$$\mathcal{W} = \{\mathbf{W}_1, \mathbf{W}_2, \cdots, \mathbf{W}_I\}$$

Time

1

• Objective function of IVA

2

Frequency

$$\mathcal{J}(\mathcal{W}) = -\mathcal{L}(\mathcal{W}) = \sum_{j=1}^{J} \sum_{n=1}^{N} \frac{G(\mathbf{y}_{j,n})}{Contrast} - 2J \sum_{i=1}^{I} \log |\det \mathbf{W}_{i}|$$

$$G(\mathbf{y}_{j,n}) = -\log \underline{p}(\mathbf{y}_{j,n}) \qquad \text{function} \qquad \mathbf{y} \text{ is a function of } \mathbf{w}$$

$$Multivariate \text{ p.d.f.} \qquad \mathbf{y}_{j,n} \qquad \mathbf{y}_{j,n} = \begin{pmatrix} y_{1j,n} \\ y_{2j,n} \\ \vdots \\ y_{Ij,n} \end{pmatrix} \begin{pmatrix} y_{1j,n} = \mathbf{w}_{1,n}^{\mathsf{H}} \mathbf{x}_{1j} \\ y_{2j,n} = \mathbf{w}_{2,n}^{\mathsf{H}} \mathbf{x}_{2j} \end{pmatrix}$$

What p.d.f. is appropriate for a spectrum vector?



Choice of multivariate p.d.f.

2

Time j

Frequency

- Necessary properties
 - Non-Gaussian (like ICA)
 - Representing higher-order dependency $y_{j,n}$ between vector components

- Well-used multivariate p.d.f.
 - Spherical super-Gaussian [Hiroe 2006], [Kim 2006]

$$p(\mathbf{y}_{j,n}) = C \exp(-||\mathbf{y}_{j,n}||_2)$$

• Time-varying Gaussian [Ono+ 2012]

$$p(\mathbf{y}_{j,n}) = \frac{C}{\sigma_{j,n}^2} \exp\left(-\frac{||\mathbf{y}_{j,n}||_2^2}{\sigma_{j,n}^2}\right)$$

Variance is time-varying. Totally it is super-Gaussian. (Show later)

Why spherical super-Gaussian?

- Let $p(\mathbf{y}) = p(y_1, y_2) = q(r)$ where $r = \sqrt{y_1^2 + y_2^2}$
- When q(r) is Gaussian, sub-Gaussian, and super-Gaussian, what dependency between y_1 and y_2 is represented?



Spherical Gaussian



In spherical Gaussian case, the value of y_1 does not change the p.d.f. of y_2 . It means y_1 and y_2 do not have any dependencies.

Spherical Sub-Gaussian



In spherical sub-Gaussian case, when $|y_1|$ is larger, $|y_2|$ tends to be smaller oppositely.

Spherical Super-Gaussian



In spherical super-Gaussian case, when $|y_1|$ is larger, $|y_2|$ tends to be also larger. Therefore, it represents co-occurrence among components.

Time-Varying Gaussian

• P.d.f. of time-varying Gaussian

$$p(\mathbf{y}_{j,n}) = \frac{C}{\sigma_{j,n}^2} \exp\left(-\frac{||\mathbf{y}_{j,n}||_2^2}{\sigma_{j,n}^2}\right)$$

- Co-occurrence among frequency components are explicitly represented by $\sigma_{j,n}^2$.
 - Shared among all frequency *i*.
 - Can be changed at each time frame *j*.



Solutions for IVA

• The objective function of IVA is also nonlinear.

$$\mathcal{J}(\mathcal{W}) = \sum_{j=1}^{J} \sum_{n=1}^{N} G(\mathbf{y}_{j,n}) - 2J \sum_{i=1}^{I} \log |\det \mathbf{W}_i|$$
$$G(\mathbf{y}_{j,n}) = ||\mathbf{y}_{j,n}||_2 = \sqrt{y_{1j,n}^2 + y_{2j,n}^2 + \dots}$$

- Similarly as ICA, three typical methods $y_{1j,n} = \mathbf{w}_{1,n}^{\mathsf{H}} \mathbf{x}_{1j}$
 - Natural gradient [Kim+ 2006, Hiroe 2006]
 - Pre-whitening + Fixed-point iteration (FastIVA) [Lee+ 2007]
 - Dr. Taesu Kim (an inventor of IVA)'s code is available https://github.com/teradepth/iva
 - Auxiliary function-based optimization (AuxIVA) [Ono2011, Ono2012b]

 $y_{2j,n} = \mathbf{w}_{2,n}^{\mathsf{H}} \mathbf{x}_{2j}$

Auxiliary Function Approach

Optimization Problem

 $J(\mathbf{\Theta}) \to \min$

Auxiliary Function

 $Q(\mathbf{\Theta}, \tilde{\mathbf{\Theta}}) \ge J(\mathbf{\Theta})$

Alternative Update Rules

Auxiliary variable update (like E-step)

$$\tilde{\boldsymbol{\Theta}}^{(k+1)} = \operatorname{argmin}_{\bar{\boldsymbol{\Theta}}} Q(\boldsymbol{\Theta}^{(k)}, \tilde{\boldsymbol{\Theta}})$$

Parameter update (like M-step)

$$\boldsymbol{\Theta}^{(k+1)} = \operatorname{argmin}_{\boldsymbol{\Theta}} Q(\boldsymbol{\Theta}, \tilde{\boldsymbol{\Theta}}^{(k+1)})$$

- Advantages
 - Stable: Convergence is guaranteed
 - Simple: No tuning parameters such as step size

But how to find useful auxiliary function is problem-dependent

More details will be explained later



Theorem for quadratic auxiliary function



Auxiliary Function for IVA



Minimizing auxiliary function

• The demixing matrix should be updated such that auxiliary function is minimized.

$$Q(\mathcal{W}, \mathbf{r}) = 2J \left[\frac{1}{2} \sum_{i=1}^{I} \sum_{n=1}^{N} \mathbf{w}_{i,n}^{\mathsf{H}} \mathbf{V}_{i,n} \mathbf{w}_{i,n} - \sum_{i=1}^{I} \log |\det \mathbf{W}_i| \right] + F(\mathbf{r})$$

$$\frac{\partial Q(\mathcal{W}, \mathbf{r})}{\partial \mathbf{w}_{i,n}} = 0 \quad \longrightarrow \quad \mathbf{w}_{i,m}^{\mathsf{H}} \mathbf{V}_{i,n} \mathbf{w}_{i,n} = \delta_{mn} \quad (m = 1, \cdots, N)$$

• From $\frac{\partial Q(\mathcal{W}, \mathbf{r})}{\partial \mathbf{w}_{i,n}} = 0$ for all $n = 1, \cdots, N$, $\mathbf{W}_{i} = \begin{bmatrix} \mathbf{w}_{i,1}^{\mathsf{H}} \\ \mathbf{w}_{i,2}^{\mathsf{H}} \\ \mathbf{w}_{i,N}^{\mathsf{H}} \end{bmatrix}$

HEAD (Hybrid Exact-Approximate Joint Diagonalization) problem [Yeredor 2009] is derived.

HEAD Problem (1/2)

For simplicity, a frequency index *i* is dropped in this slide.

Given N positive definite matrices $\mathbf{V}_1, \mathbf{V}_2, \cdots, \mathbf{V}_N$, find an N × N matrix $\mathbf{W} = (\mathbf{w}_1 \ \mathbf{w}_2 \ \cdots \ \mathbf{w}_N)^{\mathsf{H}}$ such that $\mathbf{w}_m^{\mathsf{H}} \mathbf{V}_n \mathbf{w}_n = \delta_{mn} \quad (m, n = 1, 2, \cdots, N)$

[Yeredor 2009]

ex. N=3 case

$$\mathbf{w}_{1}^{\mathsf{H}} \mathbf{V}_{1} \mathbf{w}_{1} = 1 \qquad \mathbf{w}_{1}^{\mathsf{H}} \mathbf{V}_{2} \mathbf{w}_{2} = 0 \qquad \mathbf{w}_{1}^{\mathsf{H}} \mathbf{V}_{3} \mathbf{w}_{3} = 0$$

$$\mathbf{w}_{2}^{\mathsf{H}} \mathbf{V}_{1} \mathbf{w}_{1} = 0 \qquad \mathbf{w}_{2}^{\mathsf{H}} \mathbf{V}_{2} \mathbf{w}_{2} = 1 \qquad \mathbf{w}_{2}^{\mathsf{H}} \mathbf{V}_{3} \mathbf{w}_{3} = 0$$

$$\mathbf{w}_{3}^{\mathsf{H}} \mathbf{V}_{1} \mathbf{w}_{1} = 0 \qquad \mathbf{w}_{3}^{\mathsf{H}} \mathbf{V}_{2} \mathbf{w}_{2} = 0 \qquad \mathbf{w}_{3}^{\mathsf{H}} \mathbf{V}_{3} \mathbf{w}_{3} = 1$$

HEAD Problem (2/2)

For simplicity, a frequency index *i* is dropped in this slide.

Given N positive definite matrices $\mathbf{V}_1, \mathbf{V}_2, \cdots, \mathbf{V}_N$, find an N × N matrix $\mathbf{W} = (\mathbf{w}_1 \ \mathbf{w}_2 \ \cdots \ \mathbf{w}_N)^{\mathsf{H}}$ such that $\mathbf{w}_m^{\mathsf{H}} \mathbf{V}_n \mathbf{w}_n = \delta_{mn} \quad (m, n = 1, 2, \cdots, N)$

[Yeredor 2009]

Remarks

- Number of equations = number of variables = N^2
- When N=2, it is equivalent to generalized eigenvalue problem, which can be solved in a closed form. (Show later)
- When $N \geq 3$, a closed-form solution has never been found.

Derivation of sequential update rule



Algorithm of AuxIVA



Unit vector with the *n*th element unity

Separation Example (1/6)



Separation Example (2/6)



Separation Example (3/6)



Separation Example (4/6)



Separation Example (5/6)



Separation Example (6/6)



Comparison with Natural Gradient

Three sources case



HEAD Problem in Two-Sources Case



HEAD problem is deformed to generalized eigenvalue problem, which can be solved in a closed form [Yoshioka 2008,Ono 2010]

Algorithm of Stereo AuxIVA



Comparison with Natural Gradient (2)


For simplicity, a frequency index *i* is dropped in this slide.

•
$$\mathbf{V}_2 \mathbf{u}_n = \lambda_n \mathbf{V}_1 \mathbf{u}_n \longleftrightarrow \mathbf{H} \mathbf{u}_k = \lambda_n \mathbf{u}_n \ (\mathbf{H} = \mathbf{V}_1^{-1} \mathbf{V}_2)$$

• Two eigenvectors of 2×2 matrix can be explicitly given by

$$\begin{split} \lambda_1 &= \frac{\operatorname{tr}(\mathbf{H}) + \sqrt{\operatorname{tr}(\mathbf{H})^2 - 4\operatorname{det}(\mathbf{H})}}{2} \\ \lambda_2 &= \frac{\operatorname{tr}(\mathbf{H}) - \sqrt{\operatorname{tr}(\mathbf{H})^2 - 4\operatorname{det}(\mathbf{H})}}{2} \\ \mathbf{u}_1 &= \begin{pmatrix} H_{22} - \lambda_1 \\ -H_{21} \end{pmatrix}, \quad \mathbf{u}_2 = \begin{pmatrix} -H_{12} \\ H_{11} - \lambda_2 \end{pmatrix} \end{split}$$

• See [Ono 2012b] about fast implementation using vector operation.

Implementation on iPhone

- Stereo AuxIVA was implemented on iPhone [Ono 2012b]
- Calculation time is almost linear to input signal length (RTF = 1/5 @ 16kHz, 10itr. on iPhone4)



- Demo on youtube (<u>https://www.youtube.com/watch?v=ILMbfIDMMeE</u>)
- Another Demo in Demo Session II at this ICASSP
 - DEMO-2.2: Low-Latency Real-Time Blind Source Separation for Hearing Aids
 - Time: Wednesday, April 18, 13:30 15:30
 - Location: Exhibit Hall Foyer

Tutorial structure

1. Introduction

- 1. Separation of audio/speech signals
- 2. Live demonstration

2. ICA and IVA

- 1. ICA: Independent Component Analysis
- 2. IVA: Independent Vector Analysis

3. NMF

- 1. NMF: Nonnegative Matrix Factorization
- 2. MNMF: Multichannel NMF

4. ILRMA

1. ILRMA: Independent Low-Rank Matrix Analysis

Tensor and sliced matrices



What is NMF? From "generative model" perspective





time j

What is NMF? From "generative model" perspective



time j

What is NMF? From "generative model" perspective

Matrix factorization of **X** = Restoring information about **T** and **V** from **X**



NMF as spectrogram model fitting

 Model a mixture spectrum as the sum of basis spectra scaled by time-varying magnitudes



Popular divergence measures

• Measure of difference between x and y



Geometrical understanding of NMF

- Because of the non-negativity of T, all basis vectors lie in the first quadrant.
- Because of the non-negativity of V, Tv_j can only cover the area (a convex cone) enclosed by the extended lines of all the basis vectors.
- NMF attempts to find a convex cone that is closest to all the observed vectors.



Sparsity-inducing effect of NMF

- NMF naturally produces sparse representations
 - Let $\hat{\mathbf{v}}$ be the solution to an unconstrained optimization problem: $\hat{\mathbf{v}} = \operatorname{argmin} \mathcal{D}(\mathbf{x} | \mathbf{Tv})$
 - $\mathbf{T}\hat{\mathbf{v}}$ corresponds to the closest point from \mathbf{x} in the subspace spanned by $\mathbf{t}_1, \dots, \mathbf{t}_K$.
 - Except for the case where \hat{v} is non-negative, the solution to the constrained optimization problem, $T\tilde{v}$, will be the closest point to $T\hat{v}$ in the area enclosed by the extended lines of all the basis vectors.
 - \bullet This means at least one of the elements of $\widetilde{\mathbf{v}}$ becomes 0.



Itakura-Saito divergence NMF [Févotte+2009]

 Model mixture signal as the sum of Gaussian-distributed random signals with rank-1 power spectrograms



 Maximum likelihood of T and V amounts to NMF using Itakura-Saito divergence

Local Gaussian model (LGM)

 Generative model assuming each element of a complex spectrogram to independently follow a zero-mean complex Gaussian distribution with a different variance (power)



Itakura-Saito divergence NMF (IS-NMF)

• Optimization problem: Minimize

$$\mathcal{D}_{\mathrm{IS}}(\mathbf{T}, \mathbf{V}) = \sum_{i,j} \left(\frac{|x_{ij}|^2}{\sum_k t_{ik} v_{kj}} - \log \frac{|x_{ij}|^2}{\sum_k t_{ik} v_{kj}} - 1 \right)$$
$$= \sum_{i,j} \left(\frac{|x_{ij}|^2}{\sum_k t_{ik} v_{kj}} + \log \sum_k t_{ik} v_{kj} - \cdots \right)$$

subject to $\forall i, j, k, t_{ik} \ge 0, v_{kj} \ge 0$

- How can we solve this?
 - EM algorithm [Févotte+2009]
 - Auxiliary function approach
 - -Majorization-Maximization (MM) algorithm [Kameoka+2006], [Nakano+2010], [Févotte+2011]
 - -Majorization-Equalization (ME) algorithm [Févotte+2011]

Expectation-Maximization (EM) algorithm

• When a certain likelihood function can be written as $p(\mathbf{x}|\theta) = \int p(\mathbf{x}, \mathbf{c}|\theta) d\mathbf{c}$ where **c** is a set of latent variables, a stationary point of $\log p(\mathbf{x}|\theta)$ can be found by iteratively performing the following steps:

• Expectation-Step
$$q(\mathbf{c}|\mathbf{x}) \leftarrow \frac{p(\mathbf{x}, \mathbf{c}|\theta)}{\int p(\mathbf{x}, \mathbf{c}'|\theta) d\mathbf{c}'} = p(\mathbf{c}|\mathbf{x}, \theta)$$

• Maximization-Step $\theta \leftarrow \operatorname{argmax}_{\theta} \int q(\mathbf{c}|\mathbf{x}) \log p(\mathbf{x}, \mathbf{c}|\theta) d\mathbf{c}$
 $Q(\theta, \theta') = \mathbb{E}_{\mathbf{c} \sim p(\mathbf{c}|\mathbf{x}, \theta')} [\log p(\mathbf{x}, \mathbf{c}|\theta)]$
Q function

IS-NMF optimization with EM algorithm

• Likelihood function for IS-NMF:

$$p(\mathbf{x}|\theta) = \int p(\mathbf{x}, \mathbf{c}|\theta) d\mathbf{c} = \int p(\mathbf{x}|\mathbf{c}) p(\mathbf{c}|\theta) d\mathbf{c} \quad \text{where } \theta = \{\mathbf{T}, \mathbf{V}\}$$

$$\prod_{i,j} \delta\left(x_{ij} - \sum_{k} c_{ijk}\right) \quad \prod_{k \ i,j} \prod_{i,j} \mathcal{N}_{\mathbb{C}}(c_{ijk}|0, t_{ik}v_{kj})$$

$$(independent of \theta) \quad -\log t_{ik}v_{kj} - \frac{|c_{ijk}|^2}{t_{ik}v_{kj}}$$

$$Q \text{ function} \quad -\log t_{ik}v_{kj} - \frac{|c_{ijk}|^2}{t_{ik}v_{kj}}$$

$$P(\theta, \theta') = \sum_{k} \sum_{i,j} \mathbb{E}_{c_{ijk} \sim p(c_{ijk}|x_{ij}, \theta')} \left[\log \mathcal{N}_{\mathbb{C}}(c_{ijk}|0, t_{ik}v_{kj})\right]$$

$$E-step \quad r_{ijk} \leftarrow \mathbb{E}_{c_{ijk} \sim p(c_{ijk}|x_{ij}, \theta')} \left[|c_{ijk}|^2\right] \\ = t_{ik}v_{kj} - \frac{t_{ik}^2v_{kj}^2}{\phi_{ij}} + \frac{t_{ik}^2v_{kj}^2|x_{ij}|^2}{\phi_{ij}^2} \quad \text{where } \phi_{ij} = \sum_{k} t_{ik}v_{kj}$$

$$M-step \quad \theta \leftarrow \underset{\theta}{\operatorname{argmax}} \sum_{k} \sum_{i,j} \left(-\log t_{ik}v_{kj} - \frac{r_{ikj}}{t_{ik}v_{kj}}\right)$$

Auxiliary function approach

- Techniques to find a stationary point of an objective function $D(\theta)$ using an auxiliary function $G(\theta, \alpha)$ that satisfies $D(\theta) = \min G(\theta, \alpha)$
- Majorization-minimization [Hunter & Lange 2004]
 - $\begin{aligned} \textbf{[1]} \ \alpha^{(\ell+1)} &= \operatorname*{argmin}_{\bar{\theta}} G(\theta^{(\ell)}, \alpha) \\ \textbf{[2]} \ \theta^{(\ell+1)} &= \operatorname*{argmin}_{\theta} G(\theta, \alpha^{(\ell+1)}) \\ D(\theta^{(\ell)}) &= G(\theta^{(\ell)}, \alpha^{(\ell+1)}) \\ &\geq G(\theta^{(\ell+1)}, \alpha^{(\ell+1)}) \\ &\geq D(\theta^{(\ell+1)}) \end{aligned}$
- Majorization-equalization
 [Févotte & Idier 2010]

[1]
$$\alpha^{(\ell+1)} = \operatorname*{argmin}_{\bar{\theta}} G(\theta^{(\ell)}, \alpha)$$

[2] $\theta^{(\ell+1)} \leftarrow \theta$ such that $G(\theta, \alpha^{(\ell+1)}) = G(\theta^{(\ell)}, \alpha^{(\ell+1)}), \ \theta \neq \theta^{(\ell)}$



Graphical illustration of MM and ME algorithms



Graphical illustration of MM and ME algorithms



Graphical illustration of MM and ME algorithms



• Objective: maximize
$$p(\mathbf{x}|\theta)$$
 w.r.t. θ
 $\log p(\mathbf{x}|\theta) = \log \int p(\mathbf{x}, \mathbf{c}|\theta) d\mathbf{c}$ c: latent variable
 $= \log \int q(\mathbf{c}|\mathbf{x}) \frac{p(\mathbf{x}, \mathbf{c}|\theta)}{q(\mathbf{c}|\mathbf{x})} d\mathbf{c}$ \mathbf{c} : latent variable
 $= \log \int q(\mathbf{c}|\mathbf{x}) \frac{p(\mathbf{x}, \mathbf{c}|\theta)}{q(\mathbf{c}|\mathbf{x})} d\mathbf{c}$ \mathbf{c} --- Jensen's inequality
 $\geq \int q(\mathbf{c}|\mathbf{x}) \log \frac{p(\mathbf{x}, \mathbf{c}|\theta)}{q(\mathbf{c}|\mathbf{x})} d\mathbf{c}$ Auxiliary function
E step:
Maximize auxiliary function w.r.t. q
 $q(\mathbf{c}|\mathbf{x}) \leftarrow \frac{p(\mathbf{x}, \mathbf{c}|\theta)}{\int p(\mathbf{x}, \mathbf{c}'|\theta) d\mathbf{c}'}$
 \parallel
 $p(\mathbf{c}|\mathbf{x}, \theta)$ \mathbf{c} \mathbf







• Coordinate descent of $G(\theta, \alpha)$ Contour line of $G(\theta, \alpha)$ parameter Initial parameter

 $D(\theta)$ auxiliary variable α

Motivations for auxiliary function approach

- Auxiliary function can be useful and effective when one wants to
 - handle a non-convex objective function with multiple local optima and stationary points,
 - handle an objective function that has discontinuous/non-differentiable points,
 - handle equality/inequality constraints, and
 - avoid matrix inversions.

Useful inequalities for auxiliary function design (1/7)

• Jensen's inequality for non-negative arguments



Useful inequalities for auxiliary function design (2/7)

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• Jensen's inequality for real number arguments



Useful inequalities for auxiliary function design (3/7)

1st order Taylor expansion of convex/concave functions

h: concave function $\longrightarrow h(x) \le h'(u)(x-u) + h(u)$

Equality holds when u = x

E.g.) when $h(x) = -\log x$: $\log x \le \frac{1}{u}(x-u) + \log u$ h'(u)(x-u) + h(u) h(x)

Useful inequalities for auxiliary function design (4/7)

 1st order Taylor expansion of logarithmic function Scalar case:

 $\log x \le \frac{x}{u} + \log u - 1$ Equality holds when u = x

Extension to matrix case: $\underline{\log \det X} \leq \operatorname{tr}(\mathbf{U}^{-1}\mathbf{X}) + \log \det \mathbf{U} - M$ Equality holds when $\mathbf{U} = \mathbf{X}$ Solution Equals to the sum of the logarithms of the eigenvalues of \mathbf{X}

Used in multichannel NMF frameworks [Sawada+2012, Higuchi+2014] and Positive Semidefinite Tensor Factorization [Yoshii+2013]

Useful inequalities for auxiliary function design (5/7)

• Quadratic function tangent to power functions

When
$$0 : $|x|^p \le \frac{p}{2} |u|^{p-2} x^2 + |u|^p - \frac{p}{2} |u|^p$$$

Equality holds when u = x

• Used for sparse regularization for complex NMF [Kameoka+2009]



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Useful inequalities for auxiliary function design (6/7)



• Used for sound source localization [Ono+2009][Ono+2010] and Time-domain Spectrogram Factorization (TSF) [Kameoka2015]



Useful inequalities for auxiliary function design (7/7)

Logistic function

$$(x) = \frac{1}{1 + \exp(-x)}$$

$$\geq f(\eta) \exp\left(\frac{x - \eta}{2} - \frac{\tanh(\eta/2)}{4\eta}(x^2 - \eta^2)\right)$$

Gaussian distribution function

Equality holds when $\eta = x$



IS-NMF optimization with MM algorithm [Kameoka+2006][Nakano+2010][Févotte+2011]

• Objective function to be minimized:

$$\mathcal{D}_{\mathrm{IS}}(\mathbf{T}, \mathbf{V}) = \sum_{i,j} \left(\frac{|x_{ij}|^2}{\sum_k t_{ik} v_{kj}} + \log \sum_k t_{ik} v_{kj} - \cdots \right)$$

Reciprocal function is convex in positive domain

$$\frac{1}{\sum_{k} z_{k}} = \frac{1}{\sum_{k} \lambda_{k} \frac{z_{k}}{\lambda_{k}}} \leq \sum_{k} \lambda_{k} \frac{1}{\frac{z_{k}}{\lambda_{k}}} = \sum_{k} \frac{\lambda_{k}^{2}}{z_{k}}$$
Jensen's inequality
$$\frac{1}{\sum_{k} t_{ik} v_{kj}} \leq \sum_{k} \frac{\lambda_{ijk}^{2}}{t_{ik} v_{kj}}$$

Logarithmic function is concave

$$g(u) \leq g'(u)(z-u) + g(u)$$

$$\Longrightarrow \log \sum_{k} t_{ik} v_{kj} \leq \frac{1}{u_{ij}} \left(\sum_{k} t_{ik} v_{kj} - u_{ij} \right) + \log u_{ij}$$



IS-NMF optimization with MM algorithm

[Kameoka+2006][Nakano+2010][Févotte+2011]

Majorizer

$$\mathcal{G}_{\mathrm{IS}}(\mathbf{T}, \mathbf{V}, \boldsymbol{\lambda}, \mathbf{U}) = \sum_{i,j} \left(\sum_{k} \frac{\lambda_{ijk}^2 |x_{ij}|^2}{t_{ik} v_{kj}} + \frac{1}{u_{ij}} \left(\sum_{k} t_{ik} v_{kj} - u_{ij} \right) + \log u_{ij} - \cdots \right)$$

• Update rules for λ and U

$$\lambda_{ijk} \leftarrow \frac{t_{ik} v_{kj}}{\sum_{k'} t_{ik'} v_{k'j}} \qquad \qquad u_{ij} \leftarrow \sum_{k} t_{ik} v_{kj}$$

• Update rules for T and V

$$\frac{\partial \mathcal{G}_{\mathrm{IS}}}{\partial t_{ik}} = \sum_{j} \left(-\frac{\lambda_{ijk}^2 |x_{ij}|^2}{t_{ik}^2 v_{kj}} + \frac{v_{kj}}{u_{ij}} \right) = 0 \quad \Leftrightarrow t_{ik} = \sqrt{\frac{\sum_{j} \lambda_{ijk}^2 |x_{ij}|^2 / v_{kj}}{\sum_{j} v_{kj} / u_{ij}}}$$
$$\frac{\partial \mathcal{G}_{\mathrm{IS}}}{\partial v_{kj}} = \sum_{i} \left(-\frac{\lambda_{ijk}^2 |x_{ij}|^2}{t_{ik} v_{kj}^2} + \frac{t_{ik}}{u_{ij}} \right) = 0 \quad \Leftrightarrow v_{kj} = \sqrt{\frac{\sum_{i} \lambda_{ijk}^2 |x_{ij}|^2 / t_{ik}}{\sum_{i} t_{ik} / u_{ij}}}$$

* Update rules of ME algorithm can be obtained by solving T and V that satisfies $\mathcal{G}_{\mathrm{IS}}(\mathbf{T}, \mathbf{V}, \boldsymbol{\lambda}, \mathbf{U}) = \mathcal{G}_{\mathrm{IS}}(\mathbf{T}', \mathbf{V}', \boldsymbol{\lambda}, \mathbf{U})$

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Extension to multichannel input

• Frequency-domain instantaneous mixture:



 Assume local Gaussian model (LGM) with source power spectrograms expressed using NMF (low-rank) model

 Source model assuming each element of a complex spectrogram to independently follow a zero-mean complex Gaussian distribution with a different variance (power)

$$s_{ij,n} \sim \mathcal{N}_{\mathbb{C}}(s_{ij,n} | 0, \sigma^2_{ij,n})$$
 variance

 Source model assuming each element of a complex spectrogram to independently follow a zero-mean complex Gaussian distribution with a different variance (power)



 Source model assuming each element of a complex spectrogram to independently follow a zero-mean complex Gaussian distribution with a different variance (power)

$$s_{ij,n} \sim \mathcal{N}_{\mathbb{C}}(s_{ij,n} | 0, \sigma^2_{ij,n})$$
 variance

- Allows us to incorporate power spectrogram model in $\sigma_{ij,n}^2$
- $\sigma_{ij,n}^2 = \sum_k t_{ik,n} v_{kj,n}$ corresponds to NMF model

 Source model assuming each element of a complex spectrogram to independently follow a zero-mean complex Gaussian distribution with a different variance (power)

$$s_{ij,n} \sim \mathcal{N}_{\mathbb{C}}(s_{ij,n}|0, \sigma^2_{ij,n})$$
 variance

- Allows us to incorporate power spectrogram model in $\sigma_{ij,n}^2$
- $\sigma_{ij,n}^2 = \sum_k t_{ik,n} v_{kj,n}$ corresponds to NMF model



 \mathbf{T}_n

 \mathbf{V}_n

Super-Gaussianity of local Gaussian model

• Theorem

The time average of zero-mean Gaussian distributions with time-varying variances is super-Gaussian.

$$p(x) = \sum_{n} w_n \mathcal{N}(x; 0, v_n) \qquad \left(\sum_{n} w_n = 1\right)$$

Proof omitted

Kurtosis becomes 0 if and only if all the variances are equal.

n

Using local Gaussian models implies assuming source signals to follow super-Gaussian distributions

Multichannel NMF

Probability density function of microphone array observations

$$\begin{split} \mathbf{x}_{ij} \sim \mathcal{N}_{\mathbb{C}} \left(\mathbf{x}_{ij} \middle| \mathbf{0}, \mathbf{H}_{i} \begin{bmatrix} \sigma_{ij,1}^{2} & 0 \\ 0 & \ddots & \sigma_{ij,N}^{2} \end{bmatrix} \mathbf{H}_{i}^{\mathsf{H}} \right) & \leftarrow \text{ local Gaussian model} \\ \hline \\ \frac{\log - \text{likelihood}}{\log - \text{likelihood}} & & \sum_{k} t_{ik,n} v_{kj,n} \leftarrow \text{ NMF model} \\ \mathcal{L}(\mathbf{H}, \mathbf{T}, \mathbf{V}) = \sum_{i,j} \{ -\log \det \underline{\Phi}_{ij} - \mathbf{x}_{ij}^{\mathsf{H}} \underline{\Phi}_{ij}^{-1} \mathbf{x}_{ij} \} \\ \text{where } \underline{\Phi}_{ij} = \mathbf{H}_{i} \begin{bmatrix} \sigma_{ij,1}^{2} & 0 \\ 0 & \ddots & \sigma_{ij,N}^{2} \end{bmatrix} \mathbf{H}_{i}^{\mathsf{H}} \text{ and } \sigma_{ij,n}^{2} = \sum_{k} t_{ik,n} v_{kj,n} \end{split}$$

Assumption on mixing matrix

- None [Ozerov+2010][Sawada+2012]
- → Applicable for underdetermined system
- Invertible [Kameoka+2010][Kitamura+2015]
 Specialized for determined system

Optimization

[Ozerov&Févotte2010][Kameoka+2010][Sawada+2012]

- EM algorithm [Ozerov+2010][Kameoka+2010]
- Auxiliary function approach [Sawada+2012][Kitamura+2015]

Multivariate extension of IS divergence

• Different expression of Φ_{ij}

$$\mathbf{x}_{ij} = \mathbf{H}_{i} \mathbf{s}_{ij} = \sum_{n} \mathbf{h}_{i,n} \underbrace{s_{ij,n}}_{s_{ij,n}} \sim \mathcal{N}_{\mathbb{C}}(s_{ij,n} | 0, \sum_{k} t_{ik,n} v_{kj,n}) \\ \sim \mathcal{N}_{\mathbb{C}} \left(\mathbf{x}_{ij} \Big| 0, \sum_{n} \underbrace{\mathbf{h}_{i,n} \mathbf{h}_{i,n}^{\mathsf{H}}}_{k} \sum_{k} t_{ik,n} v_{kj,n} \right) \\ \underbrace{\mathbf{H}_{i,n}}_{k} : \text{spatial property of source } n$$

 $X_{ij}(=\Phi_{ij})$

• Multivariate extension of IS divergence (a.k.a. log-determinant divergence)

$$\mathcal{L}(\mathbf{H}, \mathbf{T}, \mathbf{V}) = \sum_{i,j} \{-\log \det \hat{\mathbf{X}}_{ij} - \mathbf{x}_{ij}^{\mathsf{H}} \hat{\mathbf{X}}_{ij}^{-1} \mathbf{x}_{ij} \}$$
$$= \sum_{i,j} \{-\log \det \hat{\mathbf{X}}_{ij} - \operatorname{tr}(\hat{\mathbf{X}}_{ij}^{-1} \mathbf{x}_{ij} \mathbf{x}_{ij}^{\mathsf{H}}) \}$$

Treating NMF bases as individual sources



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Nonnegativity: from scalar to vector

	Scalar	Vector
Observation	x	$\mathbf{x} = \begin{bmatrix} x_1 \\ \vdots \\ x_M \end{bmatrix}$
Nonnegative representation	$ x ^2 = xx^*$	$X = \mathbf{x}\mathbf{x}^{H} = \begin{bmatrix} x_1 ^2 & \dots & x_1 x_M^* \\ \vdots & \ddots & \vdots \\ x_M x_1^* & \dots & x_M ^2 \end{bmatrix}$
		onnegative complex
Low-rank model	$\hat{x}_{ij} = \sum_{k=1}^{K} t_{ik} v_{kj}$	$\hat{X}_{ij} = \sum_{k=1}^{K} H_{ik} t_{ik} v_{kj}$
	 <u>Hermitian positive-semidefinite</u> matrix 	
	$H = H^H$ All eigenvalues are nonnegative	
	• Spatial property of the <i>k</i> -th NMF basis at frequency <i>i</i>	

Optimization problem of multichannel NMF

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Objective function to be minimized

$$\begin{aligned} \mathcal{L}(\mathbf{T}, \mathbf{V}, \mathbf{H}) &= \sum_{i,j} \frac{d_*(\mathsf{X}_{ij}, \hat{\mathsf{X}}_{ij})}{\bigwedge} \\ & \text{typically IS divergence} \quad \text{with } \hat{\mathsf{X}}_{ij} = \sum_{k=1}^K \mathsf{H}_{ik} t_{ik} v_{kj} \\ d_{IS}(\mathsf{X}_{ij}, \hat{\mathsf{X}}_{ij}) &= \operatorname{tr}(\mathsf{X}_{ij} \hat{\mathsf{X}}_{ij}^{-1}) - \log \det \mathsf{X}_{ij} \hat{\mathsf{X}}_{ij}^{-1} - M \end{aligned}$$



Auxiliary function design



Multiplicative update rules

Multichannel IS-NMF

IS-NMF

$$\begin{split} t_{ik} \leftarrow t_{ik} \sqrt{\frac{\sum_{j} v_{kj} \mathrm{tr}(\hat{\mathsf{X}}_{ij}^{-1} \mathsf{X}_{ij} \hat{\mathsf{X}}_{ij}^{-1} \mathsf{H}_{ik})}{\sum_{j} v_{kj} \mathrm{tr}(\hat{\mathsf{X}}_{ij}^{-1} \mathsf{H}_{ik})}}} \\ v_{kj} \leftarrow v_{kj} \sqrt{\frac{\sum_{i} t_{ik} \mathrm{tr}(\hat{\mathsf{X}}_{ij}^{-1} \mathsf{X}_{ij} \hat{\mathsf{X}}_{ij}^{-1} \mathsf{H}_{ik})}{\sum_{i} t_{ik} \mathrm{tr}(\hat{\mathsf{X}}_{ij}^{-1} \mathsf{H}_{ik})}}} \\ \mathsf{H}_{ik} \mathsf{A} \mathsf{H}_{ik} = \mathsf{B} \quad \text{Algebraic Riccati equation} \\ \mathsf{A} = \sum_{j} t_{ik} v_{kj} \hat{\mathsf{X}}_{ij}^{-1} \\ \mathsf{B} = \mathsf{H}_{ik} \left(\sum_{j} t_{ik} v_{kj} \hat{\mathsf{X}}_{ij}^{-1} \mathsf{X}_{ij} \hat{\mathsf{X}}_{ij}^{-1}\right) \mathsf{H}_{ik} \end{split}$$

$$t_{ik} \leftarrow t_{ik} \sqrt{\frac{\sum_{j} \frac{v_{kj}}{\hat{x}_{ij}} \frac{x_{ij}}{\hat{x}_{ij}}}{\sum_{j} \frac{v_{kj}}{\hat{x}_{ij}}}}}{\sum_{j} \frac{v_{kj}}{\hat{x}_{ij}}}}$$
$$v_{kj} \leftarrow v_{kj} \sqrt{\frac{\frac{\sum_{i} \frac{t_{ik}}{\hat{x}_{ij}} \frac{x_{ij}}{\hat{x}_{ij}}}{\sum_{i} \frac{t_{ik}}{\hat{x}_{ij}}}}{\sum_{i} \frac{t_{ik}}{\hat{x}_{ij}}}}}$$

Learned spatial property example

Basis-wise



The 10 bases seem to form 2 clusters, each of which corresponds to each source.



Inter-channel phase difference of source k becomes $\omega_i \frac{D \cos \theta_k}{C}$ where C is speed of sound

Clustering NMF bases for sources

• Modify the modeling of the spatial property



• \hat{X}_{ij} with source-wise spatial property

$$\hat{\mathsf{X}}_{ij} = \sum_{k=1}^{K} \sum_{n=1}^{N} z_{kn} \mathsf{H}_{in} t_{ik} v_{kj}$$

Similar multiplicative updates derived

Learned spatial property example

Basis-wise

Source-wise



Inter-channel phase difference of source k becomes $\omega_i \frac{D \cos \theta_k}{C}$ where C is speed of sound

3 music parts separation example

• 3 sources and 2 microphones (underdetermined case)



The computational burden was heavy: it took 838.30 seconds for separating 24second mixture.

4 examples in total can be found at http://www.kecl.ntt.co.jp/icl/signal/sawada/demo/mchnmf/

Determined multichannel NMF

[Kameoka+2010]

- Special case of multichannel NMF where mixing matrix H_i is invertible
- When $\mathbf{W}_i = \mathbf{H}_i^{-1}$, we can write $\mathbf{W}_i \mathbf{x}_{ij} = \mathbf{s}_{ij}$ and so

$$\begin{split} \mathcal{L}(\mathbf{H},\mathbf{T},\mathbf{V}) &= \sum_{i,j} \{-\log \det \mathbf{\Phi}_{ij} - \mathbf{x}_{ij}^{\mathsf{H}} \mathbf{\Phi}_{ij}^{-1} \mathbf{x}_{ij}\} \\ &= \sum_{i,j}^{i,j} \left(\log \det \mathbf{W}_{i}^{\mathsf{H}} \boldsymbol{\Sigma}_{ij}^{-1} \mathbf{W}_{i} - \mathbf{x}_{ij}^{\mathsf{H}} \mathbf{W}_{i}^{\mathsf{H}} \boldsymbol{\Sigma}_{ij}^{-1} \mathbf{W}_{i} \mathbf{x}_{ij}\right) \\ &= \sum_{i,j} \left\{ 2\log \det \mathbf{W}_{i} - \sum_{n} \left(\log \sigma_{ij,n}^{2} + \frac{|s_{ij,n}|^{2}}{\sigma_{ij,n}^{2}}\right) \right\} \\ &\text{where } \underline{\boldsymbol{\Sigma}_{ij}} = \begin{bmatrix} \sigma_{ij,1}^{2} & 0 \\ 0 & \ddots & \sigma_{ij,N}^{2} \end{bmatrix} \text{ and } \sigma_{ij,n}^{2} = \sum_{k} t_{ik,n} v_{kj,n} \end{split}$$

Earlier idea of Independent Low-Rank Analysis (ILRMA)

BSS methods and optimization techniques



BSS methods and optimization techniques



Categorization of LGM-based BSS methods

Generative model $\mathbf{x}_{i,j} \sim \mathcal{N}_{\mathbb{C}}(\mathbf{x}_{i,j} \boldsymbol{\mu}_{i,j}, \boldsymbol{\Phi}_{i,j})$				
Method	$oldsymbol{\mu}_{i,j}$	$oldsymbol{\Phi}_{i,j}$		
Attias (2003)	0	$\sum_{n}\sum_{j'}\sigma_{i,j-j',n}^2H_{i,j',n}$		
Ozerov & Févotte (2010)	0	$\sum_n \sigma_{i,j,n}^2 H_{i,n} + B_i$		
Duong et al. (2010)	0	$\sum_n \sigma_{i,j,n}^2 H_{i,n}$		
Kameoka et al. (2010)	$\mathbf{W}_{i}^{-1}\sum_{j'=1}^{J'-1}\mathbf{F}_{i,j'}\mathbf{x}_{i,j-j'}$	$\mathbf{W}_{i}^{-1} \mathbf{\Sigma}_{i,j} \mathbf{W}_{i}^{-1}$		
Yoshioka et al. (2011)	$\mathbf{W}_{i}^{-1} \sum_{j'=1}^{J'-1} \mathbf{F}_{i,j'} \mathbf{x}_{i,j-j'}$	$\mathbf{W}_{i}^{-1} \mathbf{\Sigma}_{i,j} \mathbf{W}_{i}^{-1}$		
Ono et al. (2012)	0	$\mathbf{W}_{i}^{-1} \mathbf{\Sigma}_{i,j} \mathbf{W}_{i}^{-1}$		
Sawada et al. (2013)	0	$\sum_n \sigma_{i,j,n}^2 H_{i,n}$		
Higuchi et al. (2014)	0	$\sum_{n}\sum_{j'}\sigma_{i,j-j',n}^2H_{i,j',n}$		
Kitamura et al. (2015)	0	$\mathbf{W}_{i}^{-1} \mathbf{\Sigma}_{i,j} \mathbf{W}_{i}^{-1}$		
López et al. (2015)	0	$\mathbf{W}_{i}^{-1} \mathbf{\Sigma}_{i,j} \mathbf{W}_{i}^{-1}$		
Adiloğlu & Vincent (2016)	0	$\sum_n \sigma_{i,j,n}^2 H_{i,n}$		
Kounades-Bastian et al. (2016)	0	$\sum_n \sigma_{i,j,n}^2 H_{i,j,n}$		

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Categorization of LGM-based BSS methods

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Constraints on $\sigma_{i,j,n}^2$

Method	$\sigma_{i,j,n}^2$	
Attias (2003)	$t_{i,k_{j,n},n}$ where $k_{j,n} \sim \pi$	
Ozerov & Févotte (2010)	$\sum_k t_{i,k,n} v_{k,j,n}$	
Duong et al. (2010)	$\sigma_{i,j,n}^2$	
Kameoka et al. (2010)	$\sum_{k,l} t_{i,k,n} a_{i,l,n} v_{k,j,n}$	
Yoshioka et al. (2011)	$v_{j,n}a_{i,j,n}$	
Ono et al. (2012)	$v_{j,n}$	
Sawada et al. (2013)	$\sum_k z_{k,n} t_{i,k} v_{k,j}$	
Higuchi et al. (2014)	$t_{i,k_{j,n},n}v_{j,n}$ where $k_{j,n} \sim \pi$	
Kitamura et al. (2015)	$\sum_k z_{k,n} t_{i,k} v_{k,j}$	
López et al. (2015)	$\sigma^2_{i,j,n}$	
Adiloğlu & Vincent (2016)	$(\sum_k t^{\mathrm{e}}_{i,k,n} v^{\mathrm{e}}_{k,j,n})(\sum_{k'} t^{\mathrm{f}}_{i,k',n} v^{\mathrm{f}}_{k',j,n})$	
Kounades-Bastian et al. (2016)	$\sum_k t_{i,k,n} v_{k,j,n}$	

Tutorial structure

1. Introduction

- 1. Separation of audio/speech signals
- 2. Live demonstration

2. ICA and IVA

- 1. ICA: Independent Component Analysis
- 2. IVA: Independent Vector Analysis

3. NMF

- 1. NMF: Nonnegative Matrix Factorization
- 2. MNMF: Multichannel NMF

4. ILRMA

1. ILRMA: Independent Low-Rank Matrix Analysis

Tensor and sliced matrices



Historical development



Historical development



Spatial and spectral models in BSS

- Spatial model: assumption of mixing/demixing system
- Spectral model: assumption for each source



- Sparse or low-rank time-freq. structure

IVA revisited: model

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- IVA extends ICA to multivariate probabilistic model [Hiroe+, 2006], [Kim+, 2006], [Kim+, 2007]
 - Frequency-domain ICA: frequency **scalar** random variables IVA: frequency **vector** random variables



IVA revisited: compared to FDICA

• Frequency-domain ICA (FDICA)



Extension of vector source model in IVA

• Frequency vector spectral model (IVA)

Co-occurrence among **frequency bins** of each source

Extend vector model to low-rank matrix model



• NMF spectral model

Co-occurrence among time-frequency slots of each source with a low-rank structure

More precise representation of time-frequency structure

Incrementation of frequency bases



ISNMF revisited: low-rank spectral model

• Itakura–Saito NMF (ISNMF) [Févotte+, 2009]

$$\mathcal{D}_{\mathrm{IS}}\left(|\mathbf{X}|^{\cdot 2} \| \mathbf{TV}\right) = \sum_{i,j} \left[\frac{|x_{ij}|^2}{\sum_k t_{ik} v_{kj}} + \log \sum_k t_{ik} v_{kj} \right]$$

Minimization of the above is equivalent to a maximum likelihood (ML) estimation with a following generative model:

At each time-frequency slot, complex-valued component x_{ij} obeys

$$\begin{aligned} x_{ij} \sim \mathcal{N}_{c} \left(x_{ij} | 0, \sum_{k} t_{ik} v_{kj} \right) &= \frac{1}{\pi \sum_{k} t_{ik} v_{kj}} \exp \left(-\frac{|x_{ij}|^2}{\sum_{k} t_{ik} v_{kj}} \right) \end{aligned}$$

If x_{ij} can be decomposed as $x_{ij} = \sum_{k} c_{ijk}$, then

 $c_{ijk} \sim \mathcal{N}_{c} \left(c_{ijk} | 0, t_{ik} v_{kj} \right)$

Parameters are also decomposed

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ISNMF revisited: low-rank spectral model



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ILRMA: unified method of IVA and NMF

- Independent low-rank matrix analysis (ILRMA) [Kitamura+, 2016] Unification of
 - 1. Estimation of demixing matrix \mathbf{W}_i (IVA spatial model)
 - 2. Low-rank approximation using $\mathbf{T}_n \mathbf{V}_n$ (ISNMF spectral model)



ILRMA: unified method of IVA and NMF

 Independent low-rank matrix analysis (ILRMA) [Kitamura+, 2016] Motivation of ILRMA?



When sources are mixed...

Cost function in ILRMA


Cost function in ILRMA

Cost function in ILRMA



Cost function in time-varying Gaussian IVA (estimates demixing matrix) Cost function in **ISNMF**

estimates demixing matrix) (estimates low-rank spectral structure)

Spatial model update: AuxIVA Spectral model update: ISNMF



All the variables are alternatively updated

Update rule of parameters in ILRMA

Maximum-likelihood-based update rules

Alternatively update both models

Spatial model (demixing matrix) Update demixing filter [Ono+, 2011] $\mathbf{V}_{i,n} = \frac{1}{J} \sum_{j} \frac{1}{r_{ij,n}} \mathbf{x}_{ij} \mathbf{x}_{ij}^{\mathrm{H}}$ $\mathbf{w}_{i,n} \leftarrow (\mathbf{W}_i \mathbf{V}_{i,n})^{-1} \mathbf{e}_n$ $\mathbf{w}_{i,n} \leftarrow \mathbf{w}_{i,n} \left(\mathbf{w}_{i,n}^{\mathrm{H}} \mathbf{V}_{i,n} \mathbf{w}_{i,n} \right)^{-\frac{1}{2}}$ Update estimated signal $y_{ij,n} = \mathbf{w}_{i,n}^{\mathrm{H}} \mathbf{x}_{ij}$

 \mathbf{e}_n : one-hot vector with one at nth element

See also (pseudo code)

Spectral model (NMF variables) Update NMF parameters [Nakano+, 2010] $\sum |u| |^2 u = (\sum t |u| |^2)^{-2}$

$$t_{ik,n} \leftarrow t_{ik,n} \sqrt{\frac{\sum_{j} |y_{ij,n}|^2 v_{kj,n} \left(\sum_{k'} t_{ik',n} v_{k'j,n}\right)}{\sum_{j} v_{kj,n} \left(\sum_{k'} t_{ik',n} v_{k'j,n}\right)^{-1}}}}{v_{kj,n} \leftarrow v_{kj,n} \sqrt{\frac{\sum_{j} |y_{ij,n}|^2 t_{ik,n} \left(\sum_{k'} t_{ik',n} v_{k'j,n}\right)^{-2}}{\sum_{j} t_{ik,n} \left(\sum_{k'} t_{ik',n} v_{k'j,n}\right)^{-1}}}}{Update estimated variance}}$$

$$r_{ij,n} = \sum_k t_{ik,n} v_{kj,n}$$

Matlab codes for ILRMA: https://github.com/d-kitamura/ILRMA

http://d-kitamura.net/pdf/misc/AlgorithmsForIndependentLowRankMatrixAnalysis.pdf

Clustering NMF bases for sources

 NMF bases should be automatically clustered into each source Introduce

$$r_{ij,n} = \sum_{k} t_{ik,n} v_{kj,n}$$

Fixed number of bases for each source





$$r_{ij,n} = \sum_{k} z_{kn} t_{ik} v_{kj}$$

Adaptive number of bases for each source



Update rule of parameters in ILRMA

• Maximum-likelihood-based update rules with Z Alternatively update both models

Spatial model (demixing matrix) Update demixing filter [Ono+, 2011] $\mathbf{V}_{i,n} = rac{1}{J} \sum_{j} rac{1}{r_{ij,n}} \mathbf{x}_{ij} \mathbf{x}_{ij}^{\mathrm{H}}$ $\mathbf{w}_{i,n} \leftarrow (\mathbf{W}_i \mathbf{V}_{i,n})^{-1} \mathbf{e}_n$ $\mathbf{w}_{i,n} \leftarrow \mathbf{w}_{i,n} \left(\mathbf{w}_{i,n}^{\mathrm{H}} \mathbf{V}_{i,n} \mathbf{w}_{i,n} \right)^{-\frac{1}{2}}$ Update estimated signal $y_{ij,n} = \mathbf{w}_{i.n}^{\mathsf{H}} \mathbf{x}_{ij}$ e_n : one-hot vector with one at *n*th element

See also (pseudo code)

http://d-kitamura.net/pdf/misc/AlgorithmsForIndependentLowRankMatrixAnalysis.pdf

Update NMF parameters [Nakano+, 2010]

$$z_{kn} \leftarrow z_{kn} \sqrt{\frac{\sum_{i,j} |y_{ij,n}|^2 t_{ik} v_{kj} \left(\sum_{k'} z_{k'n} t_{ik'} v_{k'j}\right)^{-2}}{\sum_{i,j} t_{ik} v_{kj} \left(\sum_{k'} z_{k'n} t_{ik'} v_{k'j}\right)^{-1}}}$$

$$t_{ik} \leftarrow t_{ik} \sqrt{\frac{\sum_{j,n} |y_{ij,n}|^2 z_{kn} v_{kj} \left(\sum_{k'} z_{k'n} t_{ik'} v_{k'j}\right)^{-2}}{\sum_{j,n} z_{kn} v_{kj} \left(\sum_{k'} z_{k'n} t_{ik'} v_{k'j}\right)^{-1}}}}$$

$$v_{kj} \leftarrow v_{kj} \sqrt{\frac{\sum_{i,n} |y_{ij,n}|^2 z_{kn} t_{ik} \left(\sum_{k'} z_{k'n} t_{ik'} v_{k'j}\right)^{-2}}{\sum_{i,n} z_{kn} t_{ik} \left(\sum_{k'} z_{k'n} t_{ik'} v_{k'j}\right)^{-1}}}}$$
Update estimated variance

$$r_{ij,n} = \sum_{k} z_{kn} t_{ik} v_{kj}$$

Spectral model

(NMF variables)

Historical development



Multichannel NMF revisited: model

• Multichannel NMF with full-rank covariance [Sawada+, 2013]



ILRMA and multichannel NMF

• Relationship b/w ILRMA and multichannel NMF?

Source distribution: same $p(y_{ij,n}) = \mathcal{N}_{c}(y_{ij,n}; 0, \sum_{k} t_{ik,n} v_{kj,n})$ Spatial model: different



Multichannel NMF

Mixture of full-rank covariances (and power spectrograms) $\hat{X}_{ij} = \sum_{n} \sum_{k} z_{kn} t_{ik} v_{kj} H_{i,n}$

• Rank-1 spatial model [Duong+, 2010]

Equivalent to instantaneous mixture

$$\blacksquare \quad \mathsf{H}_{i,n} = \mathbf{h}_{i,n} \mathbf{h}_{i,n}^{\mathrm{H}}$$

 $\mathbf{h}_{i,n}$: steering vector (column vector of mixing matrix) $\mathbf{H}_i = (\mathbf{h}_{i,1} \ \cdots \ \mathbf{h}_{i,N})$

ILRMA and multichannel NMF

• Multichannel NMF with rank-1 spatial model

Cost function of multichannel NMF

$$\mathcal{J} = \sum_{i,j} \left[\operatorname{tr} \left(\mathsf{X}_{ij} \hat{\mathsf{X}}_{ij}^{-1} \right) + \log \det \hat{\mathsf{X}}_{ij} \right]$$

Substitute rank-1 spatial model $H_{i,n} = h_{i,n} h_{i,n}^{H}$ into \hat{X}_{ij}

$$\hat{\mathbf{X}}_{ij} = \sum_{k} \sum_{n} \mathbf{h}_{i,n} \mathbf{h}_{i,n}^{\mathrm{H}} z_{kn} t_{ik} v_{kj}$$
$$= \sum_{n} \mathbf{h}_{i,n} \mathbf{h}_{i,n}^{\mathrm{H}} \sum_{k} z_{kn} t_{ik} v_{kj}$$
$$= \mathbf{H}_{i} \mathbf{D}_{ij} \mathbf{H}_{i}^{\mathrm{H}}$$

$$\mathbf{H}_{i} = (\mathbf{h}_{i,1} \cdots \mathbf{h}_{i,N}), \mathbf{D}_{ij} = \begin{pmatrix} \sum_{k} z_{k1} t_{ik} v_{kj} & 0 & \cdots & 0 \\ 0 & \sum_{k} z_{k2} t_{ik} v_{kj} & \ddots & \vdots \\ \vdots & \ddots & \ddots & 0 \\ 0 & \cdots & 0 & \sum_{k} z_{kN} t_{ik} v_{kj} \end{pmatrix}$$

 $X_{ij} = \sum_{k} \sum_{n} H_{i,n} z_{kn} t_{ik} v_{kj}$

ILRMA and multichannel NMF

• Multichannel NMF with rank-1 spatial model Substitute $\hat{X}_{ij} = H_i D_{ij} H_i^H$ into the cost function of multichannel NMF

$$\begin{aligned} \mathcal{J} &= \sum_{i,j} \left[\operatorname{tr} \left(\mathbf{x}_{ij} \mathbf{x}_{ij}^{\mathrm{H}} (\mathbf{H}_{i} \mathbf{D}_{ij} \mathbf{H}_{i}^{\mathrm{H}})^{-1} \right) + \log \det \mathbf{H}_{i} \mathbf{D}_{ij} \mathbf{H}_{i}^{\mathrm{H}} \right] \\ &= \sum_{i,j} \left[\operatorname{tr} \left(\mathbf{x}_{ij} \mathbf{x}_{ij}^{\mathrm{H}} \mathbf{H}_{i}^{-\mathrm{H}} \mathbf{D}_{ij}^{-1} \mathbf{H}_{i}^{-1} \right) + \log (\det \mathbf{H}_{i}) (\det \mathbf{D}_{ij}) (\det \mathbf{H}_{i})^{\mathrm{H}} \right] \end{aligned}$$

Transform the variables as $\mathbf{W}_i = \mathbf{H}^{-1}$ and $\mathbf{y}_{ij} = \mathbf{W}_i \mathbf{x}_{ij}$

$$\begin{aligned} \mathcal{J} &= \sum_{i,j} \left[\operatorname{tr} \left(\mathbf{W}_{i}^{-1} \mathbf{y}_{ij} \mathbf{y}_{ij}^{\mathrm{H}} \mathbf{W}_{i}^{-\mathrm{H}} \mathbf{W}_{i}^{\mathrm{H}} \mathbf{D}_{ij}^{-1} \mathbf{W}_{i} \right) + \log |\det \mathbf{H}_{i}|^{2} + \log \det \mathbf{D}_{ij} \right] \\ &= -2J \sum_{i} \log |\det \mathbf{W}_{i}| + \sum_{i,j} \left[\log \prod_{n} \sum_{k} z_{kn} t_{ik} v_{kj} + \operatorname{tr} \left(\mathbf{y}_{ij} \mathbf{y}_{ij}^{\mathrm{H}} \mathbf{D}_{ij}^{-1} \right) \right] \\ &= -2J \sum_{i} \log |\det \mathbf{W}_{i}| + \sum_{i,j,n} \left[\sum_{k} z_{kn} t_{ik} v_{kj} + \frac{|y_{ij,n}|^{2}}{\sum_{k} z_{kn} t_{ik} v_{kj}} \right] \right]$$
Cost in ILRMA

Summary of ILRMA

• From IVA side:

Introduce NMF spectral model (basis incrementation)

• From multichannel NMF side:

Introduce rank-1 spatial model (instantaneous mixture assumption)



Conditions

Source signals	Music signals obtained from SiSEC2011 Two microphones and two sources (determined)	
Analysis window	nalysis window 512-ms-long Hamming window	
Shift length	128 ms (1/4 shift)	
Number of bases	30 per each source/60 for all sources	
Evaluation score	Improvement ot signal-to-distortion ratio (SDR)	



Impulse response E2A (reverberation time: 300 ms)

• Two source case (ultimate nz tour)



• Conditions

Source signals	Music signals obtained from SiSEC2011 Three microphones and three sources (determined)	
Analysis window	nalysis window 512-ms-long Hamming window	
Shift length	128 ms (1/4 shift)	
Number of bases	30 per each source/90 for all sources	
Evaluation score	Improvement ot signal-to-distortion ratio (SDR)	



Impulse response E2A (reverberation time: 300 ms)



Conclusion



Advertisement: book chapters

• MNMF and ILRMA will be published from Springer in March, 2018!



Audio Source Separation (Signals and Communication Technology) 1st ed. 2018 by Ed. Shoji Makino

Ch. 5

General formulation of multichannel extensions of NMF variants;

Hirokazu Kameoka, Hiroshi Sawada, and Takuya Higuchi.

Ch. 6

Determined Blind Source Separation with Independent Low-Rank Matrix analysis; Daichi Kitamura, Nobutaka Ono, Hiroshi Sawada, Hirokazu Kameoka, and Hiroshi Saruwatari.

Thank you very much for attending this tutorial !

Hiroshi Sawada



Nobutaka Ono



Hirokazu Kameoka



Daichi Kitamura



Supplement

Supplement p.2

Scope of this tutorial

Blind Audio Source Separation

		Multi-channel 🔌 🛰		Single-channel
		over-determined $N \leq M$	under-determined $N > M$	M = 1
Utilize training data	No	ICA, IVA ILRMA	Clustering (e.g. GMM)	
		Multi-channel NMF		NMF
	Yes	D	NN-based methods	

Matlab codes for ILRMA: https://github.com/d-kitamura/ILRMA

Tutorial structure (detailed)

1. Introduction

- 1. Separation of audio/speech signals (pp. 3, 5-12)
- 2. Live demonstration (pp. 4)

2. ICA and IVA

- 1. ICA: Independent Component Analysis
 - Formulation and model (pp. 14-24)
 - Optimization methods (pp. 25-35)
 - Permutation and scaling problems (pp. 36-40)
- 2. IVA: Independent Vector Analysis
 - Formulation and source models (pp. 43-53)
 - Auxiliary function based optimization (pp. 54-62)
 - Experiments and faster algorithm in stereo case (pp. 63-74)

3. NMF

- 1. NMF: Nonnegative Matrix Factorization
 - Interpretation as generative model (pp. 77-85)

Supplement p.3

- Optimization methods (pp. 86-107)
- Expectation-Maximization algorithm (pp. 87, 88)
- Auxiliary function approach (pp. 89-107)
- Inequalities for auxiliary function design (pp. 99-105)
- 2. MNMF: Multichannel NMF
 - Formulation (pp. 110-119)
 - Auxiliary function based optimization (pp. 120-122)
 - Experiments (pp. 123-126)
 - Model categorization (pp. 127-131)

4. ILRMA

- 1. ILRMA: Independent Low-Rank Matrix Analysis
 - Developed from IVA (pp. 135-148)
 - Developed from MNMF (pp. 149-153)
 - Summary and experiments (pp. 154-158)

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- Permutation alignment [Sawada et al., 2004], [Sawada et al., 2007], [Sawada et al., 2011]
- Scaling adjustment to microphone observations [Cardoso, 1998], [Murata et al., 2001],
 - [Matsuoka and Nakashima, 2001], [Takatani et al., 2004], [Mori et al., 2006]

ICA: Independent Component Analysis

- Information-maximization approach [Bell and Sejnowski, 1995]
- Maximum likelihood (ML) estimation [Cardoso, 1997]
- Natural gradient [Amari et al., 1996], [Cichocki and Amari, 2002]
- Equivariance property [Cardoso and Souloumiac, 1996]
- FastICA [Hyvärinen et al., 2001]
- Complex-valued ICA [Bingham and Hyvärinen, 2000], [Sawada et al., 2003]
- Auxiliary function based ICA [Ono and Miyabe, 2010]

IVA: Independent Vector Analysis

- Multivariate p.d.f. [Hiroe, 2006], [Kim et al., 2006], [Kim et al., 2007]
- FastIVA [Lee et al., 2006], [Lee et al., 2007]
- HEAD problem [Yeredor, 2009]
- Auxiliary function based IVA (AuxIVA) [Ono, 2011], [Ono, 2012b], [Ikeshita et al., 2017]
- Time-varying Gaussian p.d.f. [Ono, 2012a], [Ono et al., 2012]
- Supervised or model-based IVA [Ono et al., 2012], [Lopez et al., 2015], [Nesta and Koldovský, 2017]
- Online IVA [Kim, 2010], [Taniguchi et al., 2014], [Sunohara et al., 2017]

NMF: Nonnegative Matrix Factorization

- Auxiliary function based optimization for Euclidean distance NMF and generalized Kullback-Leibler divergence NMF [Lee and Seung, 1999], [Lee and Seung, 2001]
- EM-based optimization for Itakura-Saito divergence NMF [Févotte et al., 2009]
- Auxiliary function based optimization for Itakura-Saito divergence NMF [Kameoka et al., 2006], [Nakano et al., 2010], [Févotte and Idier, 2011]
- Auxiliary function based optimization for β divergence NMF [Nakano et al., 2010], [Févotte and Idier, 2011]
- Auxiliary function based optimization for sparse NMF [Kameoka et al., 2009]

Multi-channel NMF

- EM-based optimization [Ozerov and Févotte, 2010]
- Auxiliary function based optimization [Sawada et al., 2012], [Sawada et al., 2013], [Higuchi and Kameoka, 2014]

ILRMA: Independent Low-Rank Matrix Analysis

- Earlier idea (determined multi-channel NMF) [Kameoka et al., 2010]
- Multichannel NMF with rank-1 spatial model [Kitamura et al., 2015a]
- ILRMA [Kitamura et al., 2016], [Kitamura et al., 2018]
- Relaxation of Rank-1 spatial model [Kitamura et al., 2015b]
- Maximization-equalization algorithm [Mitsui et al., 2017b]
- Optimal window length [Kitamura et al., 2017]
- Based on Student's t-distribution [Mogami et al., 2017]
- With sparse regularization [Mitsui et al., 2017a]
- With spatial regularization [Mitsui et al., 2018]

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